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INTELLIGENT MULTIMEDIA CALLS

Field of the Invention

The present invention relates to the provision of intelligent multimedia calls and in particular, though not necessarily, to the provision of intelligent video calls to user terminals making use of circuit switched access networks.

Background to the Invention

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In the following text, reference is made to certain protocols defined by the International Telecommunications Union (ITU), namely:

Protocol name	Protocol Description
H.324	Terminal for low bit-rate multimedia communication
H.223	Multiplexing protocol for low bit rate multimedia communication
H.245	Control protocol for multimedia communication
H.323	Packet-based multimedia communications systems
H.248	Gateway Control Protocol

15 There exist many services implemented over circuit switched (CS) access networks with the help of so-called Intelligent Network (IN) nodes. These services are utilized mainly with voice bearers in the CS network. The reason for this is that the devices which are utilized by IN-based services have tended to support only pure voice. There exist however multimedia protocols which are capable of utilizing the CS bearers, e.g ITU-T H.324. Implementations of these protocols can be used for pure video telephony between users which communicate with each other interactively.

A number of deficiencies of existing implementations mean that IN services cannot be used efficiently with H.324 video calls.

Many IN services used for voice calls provide for announcements and prompts 5 to calling subscribers to allow access to and selection of services by means of Dual Tone Multi-Frequency (DTMF) digits. The narrowband multimedia protocol H.324 uses an unrestricted 64kbit/s digital bearer in 3G-324M applications as seen by the IN-nodes. The content of the 64kbit/s bearer includes audio, video and control streams multiplexed using the H.223 protocol 10 (Multiplex protocol for low bit rate multimedia communication). The voice announcements played by IN-services are typically stored in PCM coding Since the prior art IN-solutions do not include the capability to format. manipulate the content of unrestricted 64kbit/s digital calls, IN services do not have the possibility to play announcements for 3G-324M calls. 15

3G-324M uses efficient audio codecs, e.g. AMR (advanced multi rate) and G.723. These codecs are not capable of transferring the DTMF tones reliably. Thus the DTMF tones are transferred in H.324 by means of control messages, i.e. the H.245 userInputIndication message. Since the prior art IN-solutions do not include the capability to detect the content of unrestricted 64kbit/s digital calls, IN services do not have the possibility to listen the DTMF tones for 3G-324M calls.

25 Summary of the Invention

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According to an aspect of the present invention there is provided a method of setting up and/or controlling a multimedia call involving an H.324 enabled user terminal and a circuit switched connection terminating at the user terminal and at a network node, the method comprising:

sending DTMF control signals over the circuit switched connection within H.245 protocol control messages; and

on the basis of said DTMF control signals, routing or re-routing the

connection to an appropriate data source or mapping the connection to an appropriate data source.

Other aspects of the present invention are set out in the appended claims.

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Brief Description of the Drawings

Figure 1 illustrates an architecture for providing multimedia services to users; Figure 2 illustrates a MG architecture of an IN node of the architecture of Figure 1;

Figure 3 illustrates an alternative architecture for providing multimedia services to users:

Figure 4 illustrates a MG architecture of a combined VSG/IN node of the architecture of Figure 3;

Figure 5 illustrates a MGC architecture of a combined VSG/IN node of the architecture of Figure 3;

Figure 6 illustrates a further embodiment of the invention in which service logic for handling DTMF control signals is contained within a service node disposed between a video gateway and a streaming server; and

20 Figure 7 is a signalling diagram showing signalling associated with the embodiment of Figure 6.

Detailed Description of Preferred Embodiments

Figure 1 illustrates a network architecture for providing video interactive services to subscribers of a mobile telecommunications service. Subscriber terminals are assumed to be terminals using the H.324 protocol for setting up and controlling multimedia communication sessions. Subscribers access the service via a GSM or UMTS circuit switched access network. Control plane signalling is carried according to the ISUP protocol.

In order to set up a multimedia session, a subscriber places a call to an IN

node. The IN node comprises a Service Control Point (SCP) and a Service Switching Point (SSP), as well as a Media Gateway (MG). The Media Gateway is illustrated in more detail in Figure 2. The SSP recognises a multimedia session setup request, and suspends set-up of the session, whilst routing information is obtained from the SCP. Calls are then directed to a Video Gateway as will be described further below, and further directed by the Video Gateway to either an H.323 network or to a packet switched server (e.g. a streaming server). The video gateway comprises a Media Gateway and a Media Gateway Controller.

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The video gateway concept is introduced to make it possible to connect circuit switched (CS) based multimedia services to packet switched (PS) based multimedia services, e.g. enabling a video call from a CS based terminal implementing the H.324 protocol to a PS based terminal implementing the H.323/SIP protocol. This video gateway is referred to as a "video interactive gateway". Currently the video gateway is known to also contain video streaming gateway (VSG) capabilities enabling a CS based terminal to connect to PS based servers. In this case, the VSG interworks between for example ISUP + H.245 and RTSP (real time streaming protocol) on the control plane, and between TDM and IP/RTP on the user plane.

Since this configuration uses normal CS bearers, e.g. 64kbit/s unrestricted and normal CS control protocols (ISUP), it can be easily integrated with IN-services as illustrated in Figure 1. The IN services platform contains capabilities to route CS calls based on various parameters, e.g. calling party location and current time. This allows called party number modifications based on these parameters. Thus the IN can forward a CS call to the VSG on the basis of various called party numbers. The VSG maps these numbers to URLs, which represent different resources, e.g. video clips provided by streaming servers.

This makes it possible to provide services such as the local weather forecast using video and audio media (e.g. based on caller's location).

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The IN services platform may contain functionality to itself initiate (without prompting) calls to one party or to several parties. This may happen for example based on time, subscriber movements, etc. Thus the IN node can make a connection between the served subscriber and a streaming server by the means described in the preceding paragraph. This makes it possible to provide push services such as wake-up calls providing business news, and advertisement videos when the served subscriber approaches a shop/restaurant.

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The IN service logic residing in the SSP receives information about the nature 10 of a call, e.g. a 3G-324M call, from the SCP with the help of ISUP and INAP/CAP (intelligent network application part/CAMEL application part) signalling. This information is used by the IN service logic to select video clips at a streaming server instead of voice announcements played by SSP/MG1. Thus the VSG is seen by the IN as an intelligent peripheral (IP). The prior art IN 15 implementation is enhanced with the H.223 de-multiplexing and multiplexing function in MG₁. The H.248 gateway control protocol already includes packages to detect H.245 messages and pass information to the MGC. This enables the IN service logic in the SSP to receive DTMF digits received in H.245 UII messages in MG1. The received DTMF digits are used by the IN 20 service to trigger appropriate actions in the service logic. This may include routing the call to different destinations, e.g. to normal video telephones or video streaming servers.

These tools allow IN-technology to create services and service groups, which can be illustrated to a user with the help of visual information. For example, the user can use the normal terminal keypad to change to another video clip/live video camera view whilst watching another video clip. The first video clip may contain audio-visual instructions about the availability of the other video clips.

The selection of the appropriate video is controlled by the IN-service by rerouting the call to another destination. This may involve usage of other service triggers, e.g. calling party location (to select e.g. the nearest camera) or

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calling party id (to select e.g. the right language).

A detailed overview of the Media Gateway (MG₁) of the IN network is shown in Figure 2. Based on information about the call type, i.e. H.324 in the H.248 Add message, and information that DTMF digit detection is requested, the controller within the MG links the H.245 handlers and H.223 multiplexers at the MG into the call, instead of linking-in the normal DTMF receiver. The H.223 multiplexer demultiplexes the H.324 user data stream into media streams and the H.245 control stream. The latter are passed to the H.245 handler. The H.245 handler contains H.245 decoder/encoder and statefull logic to control H.245 signaling. i.e:

- When H.245 userInputIndication (UII) containing the dialed digit is received from a subscriber, the digit is passed up to the H.248 handler. The mechanism to transfer detected digits in H.248 is the same as for a normal voice call. The DTMF digits are passed by the SSP to the SCP, the SCP having the service logic which determines the appropriate actions.
- The H.245 handler gathers data relevant to the MG from the H.245 signaling between end-points, e.g. H.223 mux configuration data is transferred in an H.245 MultiplexentrySend (MES) message in order to ensure that both endpoints utilise the same multiplexing algorithm.
- H.245 signaling is triggered and suppressed when needed, e.g. if IN-services
 decides to reroute the call to a new destination whilst media channels are
 open to an old destination, the H.245 handler must close the channels to the
 calling party prior to opening channels between the new destination and the
 calling party.

One problem with the approach described above is that separated IN and VSG architectures may create delays, eg. H.223 multiplexing in MG₁. This problem may be solved by combining the IN and VSG into a single node. This makes it possible for the IN to utilize the same resources and the same H.223 demultiplexer as the VSG, which are needed for VSG functions. This approach is illustrated in Figure 3.

A detailed overview of the MG of the combined IN network/video gateway node is shown in Figure 4. The architecture is the same as in the normal MG of the VIG/VSG. The H.223 multiplexer within the MG demultiplexes the H.324 user data stream between media streams and the H.245 control stream. Deultiplexing happens according to the configuration parameters received from MGC. Media streams are transferred to RTP handlers and further to an IP network. The H.245 control stream including UII message is transferred via H.248 to MGC. All this happens in the same way as in the normal VIG/VSG.

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A detailed overview of the SSP/MGC of the combined IN network/video gateway node is shown in Figure 5. An H245 handler within the MGC decodes the H245 messages and extracts the H245 UII. These are passed via the a controller and INAP stack from the MGC to the SCP, which processes them as standard DTMF digits using the applicable service logic. The Controller within the SSP/MGC contains logic to reroute the call based on the commands received from the SCP, via INAP. This involves establishing a new RTSP session. Depending on the parameters on stream content (Session Descriptor Protocol, SDP), the controller may need to reopen the channels towards the calling party with new codec parameters with the help of the H.245 handler and modify the codec parameters in the MG.

The architectures described here allow the IN and Internet streaming servers to be combined with the help of the VSG, to provide multimedia intelligence.

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Figure 6 illustrates a further embodiment of the present invention and which is implemented without the need for an Intelligent Network. In this embodiment, the relevant service logic previously contained in the SSP and SCP is incorporated into the MGC of the video gateway. Using the Simple Network Management Protocol (SNMP), the operation of the MGC is controlled by an Operation and Maintenance (O&M) system.

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In a modification to this embodiment, for the streaming gateway scenario, the service logic may be decoupled from the video gateway, and relocated to a node located between the video gateway and the streaming servers. The signalling diagram of Figure 7 illustrates signalling associated with this modified embodiment. The new service node supports RTSP. In this embodiment, DTMF signals are extracted from the H.245 control messages at the MGC of the video gateway, and are forwarded to the new service node as "X-VIG-DTMF" elements of the RTSP message "Set Parameter". The service node then identifies the appropriate addresses (URLs) of the packet switched servers using the DTMF signals.

This modification may also be applied in other scenarios apart from that of video streaming. Consider for example the following scenarios:

- 1) A video gateway may be used for 3G-subscribers (using 3G-324M terminals) to connect to IP-based video mail systems. In this scenario the 3G-subscriber can use DTMF signals (i.e. transported as H.245 UII) to control video mail operations, e.g. to watch the next video message. The MGC within the video gateway communicates with a video mail system using H323, which includes H.245, i.e. the video gateway is transparent to H.245 UII messages.
- 2) The video gateway may interconnect H.324 terminals to the IMS (IP multimedia system, as defined by 3GPP). In this scenario the H.324 subscriber could use DTMF to control IMS services. Thus the actual call could be truly interactive video end-to-end and DTMF could be used for example to control a multiparty video conference (provided by IMS).

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It will be appreciated by the person of skill in the art that various modifications may be made to the above described embodiments without departing from the scope of the present invention. For example, the invention may be implemented without the need for a video gateway in the communication path.

This might arise when the IN controls calls between two H.324 terminals.

Abbreviations used in the preceding description include:

AMR	Adaptive Multi-Rate codec
CAMEL	Customized Applications for Mobile network Enhanced Logic
CAP	CAMEL Application Part
CS	Circuit Switched
DTMF	Dual Tone Multi-Frequency
GSM	Global System for Mobile communications
IMS	IP multimedia system
IN	Intelligent Network
INAP	Intelligent Network Application Part
IP	Internet Protocol
ISUP	ISDN User Part
MG	Media Gateway
MGC	Media Gateway Controller
PCM	Pulse Code Modulation
PS	Packet Switched
RTSP	Real Time Streaming Protocol
SCP	Service Control Point
SNMP	Simple Network Management Protocol
SSP	Service Switching Point
TDM	Time Division Multiplexed
UMTS	Universal Mobile Telecommunication service
URL	Universal Resource Locator
VIG	Video Interactive Gateway
VSG	Video Streaming Gateway